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Applicant: **N.V. Philips' Gloeilampenfabrieken**
Groenewoudseweg 1
NL-5621 BA Eindhoven(NL)

Inventor: **Bergmans, Johannes Wilhelmus**
Maria
c/o INT. OCTROOIBUREAU B.V. Prof.
Holstlaan 6
NL-5656 AA Eindhoven(NL)

Representative: **Van den Brom, Arend**
Albertus et al
INTERNATIONAAL OCTROOIBUREAU B.V.
Prof. Holstlaan 6
NL-5656 AA Eindhoven(NL)

Data transmission system comprising a decision feedback equalizer and using partial-response techniques.

A system for transmitting an n -level data signal (d_k) at a given symbol rate $1/T$ comprises a data transmitter (1), a transmission channel (2) and a data receiver (3) with an equalizer (30) of the decision feedback type. By arranging this equalizer (30) for forming an estimate (\tilde{c}_k) of a virtual m -level data signal (c_k) instead of the output signal (b_k) of the data transmitter (1) error propagation in the equalizer (30) is considerably reduced without thereby appreciably adding to the implementation-complexity of the system.

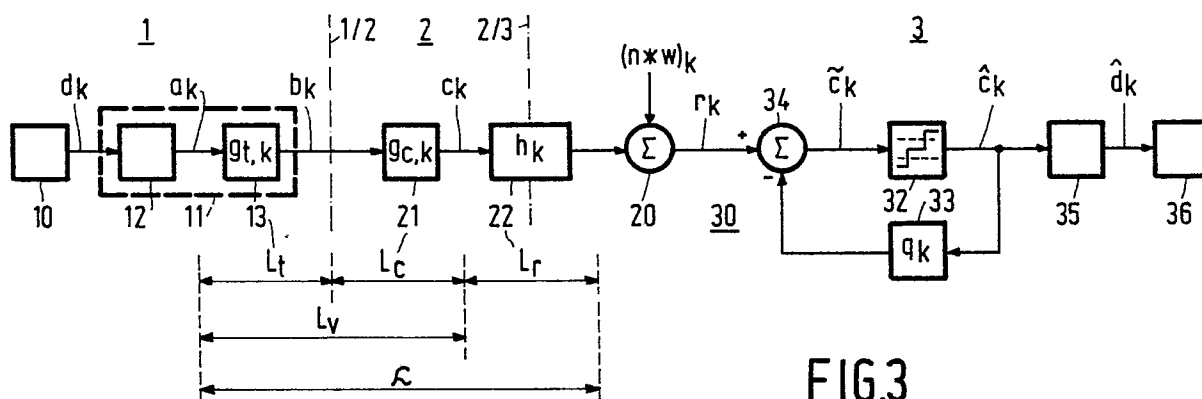


FIG.3

Data transmission system comprising a decision feedback equalizer and using partial-response techniques.

The invention relates to a system for transmitting an n-level data signal at a given symbol rate $1/T$, this system being constituted by a data transmitter with a data signal source, a transmission channel and a data receiver with an equalizer of the decision feedback type, which equalizer comprises a feedforward filter connected between the input of the data receiver and a first input of a difference circuit, a symbol decision circuit connected to the output of the difference circuit, and a feedback filter connected between the output of the symbol decision circuit and a second input of the difference circuit, in which the linear part of the transmission path between the output of the data signal source and the first input of the difference circuit can be described by a linear signal transformation L .

Such a system is generally known and is described, for example, in the book "Digital Communications" by J.G. Proakis, McGraw-Hill, 1983, Chapter 6, Section 6.5, pp. 382-386. In such systems the feedforward filter belonging to the equalizer provides for suppression of noise and cancellation of pre-cursive intersymbol interference (ISI), whilst post-cursive intersymbol interference (ISI) is cancelled with the aid of the feedback filter synthesizing a replica of this interference on the basis of the symbol decisions already formed, by which replica is subtracted from the output signal of the feedforward filter. In the system known from the book by Proakis the equalizer is arranged for forming at the input of the symbol decision circuit an estimate of a data signal generated by the data transmitter. Normally, this estimate relates to the original n-level data signal, but in the case when the data transmitter includes a linear encoder, it is likewise possible to have this estimate relate to the output signal of the encoder and reconstruct in the data receiver the original n-level data signal from the symbol decisions formed then. The latter possibility occurs, for example, in ISDN transmission systems in which pseudo-ternary transmission codes are used, compare the article "A Baud-Rate Line-Interface for Two-Wire High-Speed Digital Subscriber Loops" by C.A. Ehrenbard and M.F. Tompsett, Proc. GLOBECOM 1982, Miami, USA, pp. D.8.4.1 - D.8.4.5, in which the use of a bipolar transmission code is described.

In strongly dispersive transmission channels the output signal of the feedforward filter shows a strongly post-cursive intersymbol interference (ISI). Since the feedback filter has to synthesize a replica of this post-cursive ISI, erroneous symbol decisions applied to the feedback filter will more seriously affect subsequent symbol decisions according as the transmission channel is more dispersive. This undesired continuing influence of erroneous symbol decisions is known as error propagation and entails a degradation of the transmission quality, as appears, for example, from Fig. 6.5.2 on page 386 of the above book by Proakis.

The invention has for its object to provide a novel concept of a system of the type set forth in the preamble in which the said error propagation is reduced considerably without appreciably adding to the implementation-complexity of the system.

Thereto, a system according to the invention is characterized in that the equalizer is arranged for forming at the input of the symbol decision circuit an estimate of a virtual m-level data signal correlating with the n-level data signal applied to the input of the linear part of the transmission path according to a linear signal transformation L_v which substantially characterizes the linear signal transformation L and corresponds with a partial-response polynomial $g_v(D)$ with D being a delay operator representing the symbol interval T .

For completeness it should be observed that the m-level data signal to be estimated is virtual if and only if $g_v(D) \neq 1$, and if also $g_v(D) \neq g_t(D)$, where $g_t(D)$ is the partial-response polynomial corresponding with a linear signal transformation L_t optionally performed in the data transmitter.

The post-cursive ISI in the output signal of the feedforward filter is substantially described by the linear signal transformation L_v . According to the partial-response technique which is used in conformity with the novel concept, the major part of this ISI may be considered to be controlled desired ISI, so that only a small amount of undesired residual ISI remains which has to be cancelled by the feedback filter. The achieved reduction of the amplitude of the feedback filter output signal results in that erroneous symbol decisions applied to the feedback filter have a less strong influence on subsequent symbol decisions, thereby achieving the intended reduction in error propagation.

An embodiment of the system according to the invention that is attractive with respect to its implementation is characterized in that the data transmitter comprises a precoder connected between the data signal source and the input to the linear part of the transmission path for performing a non-linear signal transformation NL_v which is unambiguously determined by the linear signal transformation L_v , in conformity with the partial-response technique, and in that the feedback filter in the data receiver is connected to the output of the symbol decision circuit through a decoder and a precoder which is identical with the precoder

in the data transmitter, said decoder performing a memoryless inverse signal transformation L_V^{-1} •
 NL_V^{-1} which converts the m-level symbol decisions into an n-level data signal corresponding with the
 original n-level data signal. The precoder connected to the decoder subsequently converts this n-level data
 signal into a replica of the precoded n-level data signal generated in the data transmitter applied to the input
 5 to the linear part of the transmission path. In this way the condition generally to be imposed on decision
 feedback equalization that the input signal of the feedback filter is linearly related to the output signal of the
 feedforward filter is satisfied. Besides, an n-level data signal is applied to the feedback filter and because n
 is smaller than m , a digital implementation of this filter is thus simpler than when the formed m-level symbol
 decisions are applied directly.

10 A further advantage of this embodiment is the possibility of adaptively adjusting the feedback filter and
 also the feedforward filter in the data receiver of the system under control of an error signal which can be
 simply obtained and is representative of the difference between the input signal of the symbol decision
 circuit and a symbol that can be derived from the input signal of the feedback filter by performing the linear
 signal transformation L_V .

15 This adaptive embodiment finally enables to further improve the already achieved transmission quality
 by adding a relatively simple non-adaptive post-detector to which the input signal of the symbol decision
 circuit is applied.

The invention will be further explained hereinbelow with reference to the drawing, in which:

Fig. 1 shows a block diagram of a conceptual embodiment of a data transmission system in which
 20 the invention can be used;

Fig. 2 shows a functional discrete-time model of the system of Fig. 1 when conventional measures
 are employed;

Fig. 3 shows a functional discrete-time model of the system of Fig. 1 when the measures according
 to the invention are employed;

25 Fig. 4 shows a functional discrete-time model of an attractive embodiment of a system according to
 the invention; and

Fig. 5 shows a functional discrete-time model of an adaptive embodiment of a receiver of a system
 according to the invention.

In Fig. 1 a block diagram is shown of a system for data signal transmission with a data transmitter 1, a
 30 transmission channel 2 and a data receiver 3. The data transmitter 1 comprises a data signal source 10 for
 generating a data signal. This data signal is converted by an encoder 11 into a data signal which is
 transmitted through transmission channel 2 at a symbol rate $1/T$. The intersymbol interference (ISI) and
 noise developed during this transmission are combated in the data receiver 3. Thereto, data receiver 3
 comprises an equalizer 30 of the decision feedback type which includes a feedforward filter 31 which is
 35 dimensioned for suppressing in the best way possible pre-cursive ISI and noise. On the basis of symbol
 decisions which are formed in a symbol decision circuit 32 a feedback filter 33 subsequently forms a
 cancelling signal for post-cursive ISI which is subtracted from the output signal of feedforward filter 31 by
 means of a difference circuit 34 for obtaining the input signal of symbol decision circuit 32. Finally, from the
 formed symbol decisions a decoder 35 forms a replica of the original data signal which is applied to a data
 40 signal sink 36.

To illustrate the problem for which the invention provides a solution, Fig. 2 shows a functional discrete-
 time model of the system of Fig. 1 when employing conventional measures. In the Figs. 1 and 2
 corresponding elements are denoted by the same reference symbols. The model of Fig. 2 is given for the
 case in which data signal source 10 generates a binary data signal and data transmitter 1 applies a ternary
 45 data signal to transmission channel 2.

A binary data signal d_k generated by data signal source 10 is converted by a non-linear part 12 of the
 encoder 11 into a likewise binary data signal a_k which, subsequently, by the linear part 13 of the encoder 11
 is converted into a ternary data signal b_k to be applied to discrete-time transmission channel 2. To
 characterize the operation performed in this linear part 13 a partial-response polynomial $g_t(D)$ can be used,
 50 D being a delay operator representing the symbol interval T . Further details about these partial-response
 polynomials are to be found, for example, in the article "Partial-Response Signaling" by P. Kabal and S.
 Pasupathy, IEEE trans. Commun., Vol. COM-23, No. 9, pp. 921-934, September 1975. For explaining the
 following description it should be observed that such polynomials generally have a relatively low order and
 also, apart from an otherwise unimportant scale factor, only have integral-valued coefficients. In the present
 55 case, for the purpose of illustration, the bipolar response $1-D$ for the polynomial $g_t(D)$ is chosen such that

$$b_k = a_k - a_{k-1}. \quad (1)$$

The ternary data signal b_k is converted into an output signal r_k by the cascade arrangement of
 transmission channel 2 and feedforward filter 31 in Fig. 1 according to

$$r_k = (b * (f * w))_k + (n * w)_k, \quad (2)$$

where the symbol "*" denotes the linear convolution-operator, f_k and w_k represent the discrete-time impulse responses of transmission channel 2 and feedforward filter 31, respectively, and n_k represents an additive discrete-time noise signal which is added by means of a summator 20.

5 With a proper dimensioning of the feedforward filter 31 of Fig. 1 it holds that the signal r_k contains virtually only post-cursive ISI. This implies that $(f * w)_k$ can significantly differ from zero only for non-negative instants k . In the present system post-cursive ISI is combated by making feedback filter 33 have a causal impulse response p_k for which holds

10

$$p_k = \begin{cases} 0, & k \leq 0, \\ (f * w)_k, & k \geq 1, \end{cases} \quad (3)$$

15

and applying to this feedback filter 33 the symbol decisions \hat{b}_k which are formed by decision circuit 32. As a result of the causal character of feedback filter 33 its output signal is at any instant k only determined by symbol decisions \hat{b}_{k-i} with $i \geq 1$ that have already been formed. Under normal operating conditions these symbol decisions are correct, so that the output signal of the feedback filter 33 can be described as

$$(\hat{b} * p)_k = (b * p)_k. \quad (4)$$

20 The output signal \hat{b}_k of difference circuit 34 can now be described as

$$\hat{b}_k = r_k - (\hat{b} * p)_k. \quad (5)$$

In the case in which signal r_k only contains post-cursive ISI, this formula when utilizing formulas (2), (3) and (4) can be simplified to

$$\hat{b}_k = b_k + (n * w)_k = b_k + n_k, \quad (6)$$

25 where n_k represents the version of noise signal n_k that is attenuated in amplitude by feedforward filter 31. According to the latter formula, in the absence of error propagation, at the input of symbol decision circuit 32 an ISI-free estimate \hat{b}_k is formed of the data signal b_k at the output of data transmitter 1.

For strongly dispersive transmission channels 2 the output signal of feedforward filter 31 usually shows a strongly postcursive ISI because the impulse response $(f * w)_k$ for $k \geq 1$ significantly differs from zero. Consequently, the impulse response p_k of feedback filter 33 according to formula (3) will also assume values significantly differing from zero for $k \geq 1$. This will unavoidably cause a relatively large effect of erroneous symbol decisions \hat{b}_{k-1} with $i \geq 1$ that have already been formed on symbol decisions \hat{b}_{k+i} with $i \geq 0$ that still have to be formed.

35 In Fig. 2 the cascade arrangement of the linear part 13 of encoder 11 in data transmitter 1 the transmission channel 2 and the feedforward filter 31 of equalizer 30 in data receiver 3 constitutes the linear part of the transmission path between the output of data signal source 10 and the first input of difference circuit 34. The operation of this cascade arrangement (13,2,31) can be described by a linear signal transformation \mathcal{L} , as is symbolically shown in Fig. 2. Instead of inserting summator 20 at the input of feedforward filter 31 in this cascade arrangement (13,2,31) it is equally possible to insert same at the output of this feedforward filter 31 having impulse response w_k . On the basis of the above considerations it will be evident that in the latter case summator 20 has to add to the output signal of this cascade arrangement (13,2,31) an additive noise signal $(n * w)_k$ in order to produce the same signal r_k at the first input of difference circuit 34 as in the case shown in Fig. 2.

45 The latter option is used for elucidating the description of Fig. 3 showing a functional discrete-time model of the system of Fig. 1 when utilizing the measures according to the invention. In the Figs. 1, 2 and 3 corresponding elements are denoted by the same reference symbols.

The linear part 13 of encoder 11 in Fig. 3 again performs an operation which is characterized by the partial-response polynomial $g_t(D) = 1 - D$. At the output of data transmitter 1 in Fig. 3 then again a ternary data signal b_k according to formula (1) will occur

$$b_k = a_k - a_{k-1} \quad (7)$$

50 and at the first input of difference circuit 34 in data receiver 3 a signal r_k according to formula (2)

$$r_k = (b * (f * w))_k + (n * w)_k. \quad (8)$$

In many cases it is possible to present a relatively simple partial-response polynomial $g_c(D)$ such that the associated impulse response $g_{c,k}$ - which is built up out of the respective coefficients of the polynomial - forms a proper styling of the impulse response $(f * w)_k$ of the cascade arrangement of transmission channel 2 and feedforward filter 31. This implies that the linear signal transformation corresponding with the impulse response $(f * w)_k$ representative of the overall linear transmission distortion can be considered to be built up

as a sequence of partial-response transformation L_c which corresponds with $g_c(D)$, and a residual transformation L_r which takes into account the generally minor effect of the residual linear transmission distortion. In the present example, the duobinary response $1 + D$ is taken for $g_c(D)$, which response is illustrative of many transmission channels 2 having a low-pass character such as, for example, ISDN connections in the local public telephone network. This conceptual splitting-up is expressed in Fig. 3 by a partial-response circuit 21 corresponding with linear signal transformation L_c and having an impulse response $g_{c,k}$, which circuit 21 is followed by a residual circuit 22 corresponding with linear signal transformation L_r and having an impulse response h_k . In partial-response circuit 21 ternary data signal b_k at the output of data transmitter 1 is converted into a virtual m-level data signal c_k (signal c_k is a "virtual" signal because it is not explicitly visible at any point between the in and output of the physical transmission channel 2). Then, for this m-level data signal c_k it holds that

$$c_k = (b * g_c)_k, \quad (9)$$

which expression for the assumed duobinary response $g_c(D) = 1 + D$ is simplified to

$$c_k = b_k + b_{k-1}. \quad (10)$$

On the basis of formula (7) it then follows that c_k is related to binary data signal a_k at the input of linear part 13 of encoder 11 in data transmitter 1 according to

$$c_k = a_k - a_{k-2}, \quad (11)$$

so that c_k in this case is a ternary data signal (thus $m = 3$). This relationship can be described by a linear signal transformation L_v which can be assumed to be built up as a sequence of partial-response transformations L_t and L_c which correspond with the polynomials $g_t(D)$ and $g_c(D)$, as represented in Fig. 3. The signal transformation L_v then corresponds with a partial-response polynomial $g_v(D)$ for which holds

$$g_v(D) = g_t(D) \cdot g_c(D). \quad (12)$$

In the present example the bipolar response $1 - D$ is chosen for $g_t(D)$ and the duobinary response $1 + D$ for $g_c(D)$, so that

$$g_v(D) = (1-D)(1+D) = 1-D^2. \quad (13)$$

In view of the generally relatively small residual transmission distortion which is represented by the impulse response h_k , the signal transformation L of the linear part (13,2,31) of the transmission path between the output of signal source 10 and the first input of difference circuit 34 is substantially characterized by the linear signal transformation L_v which is performed by the cascade arrangement of linear part 13 of encoder 11 and partial-response circuit 21.

The described conceptual splitting-up becomes explicitly visible in data receiver 3 of Fig. 3 because in accordance with the invention equalizer 30 is arranged for forming at the input of symbol decision circuit 32 an estimate \tilde{c}_k of the virtual data signal c_k instead of an estimate \tilde{b}_k of the data signal b_k at the output of data transmitter 1. The task to be performed by the equalizer 30 is less exacting in the case of Fig. 3 in view of the relatively small residual transmission distortion which is represented by the impulse response h_k . This can be shown by a further analysis of the model of Fig. 3. As appears from the splitting-up of Fig. 3 the signal r_k at the first input of difference circuit 34 can be written as

$$r_k = (c * h)_k + (n * w)_k. \quad (14)$$

By analogy with the foregoing, under normal operational conditions the already formed symbol decisions \hat{c}_{k-i} with $i \geq 1$ may be assumed to be correct. Applying these correct symbol decisions to feedback filter 33, now having an impulse response q_k , then results in an output signal

$$(\hat{c} * q)_k = (c * q)_k. \quad (15)$$

By utilizing formulas (14) and (15) it now appears that at the input of symbol decision circuit 32 a signal \tilde{c}_k develops having the form

$$c_k = (c * h)_k - (c * q)_k + (n * w)_k. \quad (16)$$

In order to let this signal \tilde{c}_k be as good an approximation as possible of the virtual data signal c_k , it is necessary according to this formula that the impulse response q_k of feedback filter 33 be a faithful copy of the causal part of the impulse response h_k , that is to say

$$q_k = \begin{cases} 0, & k \leq 0, \\ h_k & k \geq 1, \end{cases} \quad (17)$$

As appears from the foregoing, impulse response h_k usually represents only a small amount of linear transmission distortion, so that the impulse response q_k will take on relatively small values, and already formed erroneous symbol decisions \hat{c}_{k-i} with $i \geq 1$ only affect to a limited extent the symbol decisions \hat{c}_{k+i} with $i \geq 0$ still to be formed.

The reduction of error propagation achieved thus can be aptly illustrated with reference to the situation in which no residual linear transmission distortion occurs, so that

$$h_k = \delta_k, \quad (18)$$

where δ_k represents the Kronecker delta function. The linear signal distortion introduced by the cascade arrangement of transmission channel 2 and feedforward filter 31 can then be characterized exactly in both Fig. 2 and Fig. 3 by the partial-response transformation L_c , so that

$$(f^*w)_k = g_{c,k}. \quad (19)$$

According to the conventional approximation of Fig. 2 the impulse response p_k according to formula (3) is a replica of the part with $k \geq 1$ of $(f^*w)_k$, that is to say

$$p_k = \begin{cases} 0, & k \leq 0, \\ g_{c,k}, & k \geq 1, \end{cases} \quad (20)$$

For the chosen duobinary response $g_c(D) = 1 + D$ it holds that $g_{c,1} = 1$ and $g_{c,k} = 0$ for $k \geq 2$, so that the first coefficient of the feedback filter 33 has a large non-zero value which may lead to significant error propagation. Conversely, the approximation according to the invention results in a feedback filter 33 whose impulse response g_k is a replica of the part with $k \geq 1$ of the impulse response h_k , which part according to formula (18) is equal to zero for all $k \geq 1$. Consequently, all coefficients of feedback filter 33 are also equal to zero, so that error propagation is eliminated completely. It will be evident that this ideal situation, in which a feedback filter 33 is actually redundant, will not occur in practice. However, in general it will still hold that the first coefficients q_k according to Fig. 3 then have a considerably smaller amplitude than the corresponding first coefficients p_k according to Fig. 2, so that error propagation is accordingly smaller.

In the configuration as shown in Fig. 3 an m -level signal c_k is applied to feedback filter 33, where $m = 3$ for the present example with $g_v(D) = 1 - D^2$. By carrying out in encoder 11 of data transmitter 1 a suitable non-linear signal transformation NL_v , it is possible to reduce this number of m signal levels and thus simplify a digital implementation of feedback filter 33.

This possibility is represented in Fig. 4 showing a functional discrete-time model of a system according to the invention. In the Figs. 3 and 4 corresponding elements are denoted by the same reference symbols.

In addition to the said non-linear signal transformation NL_v other non-linear signal processes too can generally take place in the non-linear part 12 of encoder 11. To simplify the following description these other non-linear signal processes are assumed to be incorporated in data signal source 10.

As explained hereinbefore, the operation of equalizer 30 according to the invention is aimed towards combating the residual linear transmission distortion which is represented by the impulse response h_k . Consequently, with a proper functioning of equalizer 30 the relation between the data signal a_k at the input of the linear part (13,21,22) of the transmission path and the input signal \tilde{c}_k of symbol decision circuit 32 can also be characterized by the linear signal transformation L_v . Since this linear signal transformation L_v in its turn is characterized by a partial-response polynomial $g_v(D)$, according to the said article by Kabal and Pasupathy there is a non-linear signal transformation NL_v denoted "precoding" and having the feature that the sequence of the inverse operations L_v^{-1} and NL_v^{-1} of L_v and NL_v , respectively, is a simple memoryless inverse signal mapping (MIM) which can be symbolically denoted $L_v^{-1} \bullet NL_v^{-1}$. By using this precoding NL_v in the non-linear part 12 of encoder 11 it is achieved that from the formed symbol decisions \hat{c}_k a direct estimate \hat{d}_k of input signal d_k of encoder 11 can be obtained by carrying out this memoryless inverse signal mapping MIM in decoder 35. By applying the data signal \hat{d}_k obtained thus to a precoder 37 which is identical with precoder 12 in data transmitter 1, an estimate \hat{a}_k is obtained of data signal a_k at the input of the linear part (13,21,22) of the transmission path and this estimate \hat{a}_k is applied to feedback filter 33. Thus, the condition generally to be imposed on the decision feedback equalization that the input signal of feedback filter 33 be linearly related to the signal at the first input of difference circuit 34 is satisfied. Since the precoded data signal \hat{a}_k has the same number of n amplitude levels as the original data signal d_k , a digital implementation of feedback filter 33 is simpler in Fig. 4 than in Fig. 3, in which a data signal with $m > n$ amplitude levels is applied to feedback filter 33. In the present example with $g_v(D) = 1 - D^2$ not a ternary, but a binary data signal is applied to feedback filter 33.

As appears from the foregoing, there is a relationship between the data signals c_k and a_k that can be characterized by the linear signal transformation L_v . Therefore in absence of erroneous symbol decisions \hat{c}_k the same holds for the relationship between the data signals \hat{c}_k and \hat{a}_k of Fig. 4. Expressed in a formula this means that

$$\hat{c}_k = (\hat{a} * g_v)_k. \quad (21)$$

In order to realize the same output signal of the feedback filter 33 in the configuration of Fig. 4 as in Fig. 3, feedback filter 33 in Fig. 4 has to have an impulse response q_k^1 , so that

$$(\hat{a} * q^1)_k = (\hat{c} * g)_k. \quad (22)$$

On the basis of the relationship between the data signals \hat{c}_k and \hat{a}_k according to formula (21), q_k^1 has to be related to q_k according to formula (22) as

$$q_k^1 = (q * g_v)_k. \quad (23)$$

The convolution in formula (23) generally has a shortening effect on the impulse response of feedback filter 33, as will now be explained.

In the absence of erroneous symbol decisions data signal \hat{c}_k at the output of symbol decision circuit 32 has a controlled ISI structure which is characterized by the linear signal transformation L_v . For the prevailing partial-response transformations L_v this structure leads to zeros in the amplitude spectrum of data signal \hat{c}_k , which zeros are often situated at the frequency 0 and/or at the Nyquist frequency $1/(2T)$.

As the above has shown, feedback filter 33 should cancel a residual transmission distortion which is represented by the impulse response h_k . The desired feedback filter output signal defined well in this manner has to be generated in Fig 3 by a convolution of data signal \hat{c}_k at its input and its impulse response q_k . As the amplitude spectrum of this input signal \hat{c}_k has zeros at frequencies determined by L_v , the transfer function of feedback filter 33 around these frequencies can be chosen freely without an appreciable effect on the desired output signal. Especially with an adaptive adjustment of feedback filter 33 as shown in Fig. 3 this freedom may inadvertently result in feedback filter 33 having a large transfer at the said frequencies determined by L_v . Such a large transfer is attended with an impulse response q_k of feedback filter 33 which extends over a large time span and/or has large amplitude values, and thus may lead to serious error propagation in both cases. According to formula (23) impulse response q_k^1 of feedback filter 33 in Fig. 4 is determined by the convolution of the impulse response $g_{v,k}$, which itself is determined by the linear signal transformation L_v , and the just described impulse response q_k of feedback filter 33 in Fig 3. Thus, it is achieved that a possible large transfer of feedback filter 33 in Fig. 3 at the said frequencies determined by L_v is cancelled completely or substantially completely in Fig. 4 by the very small transfer at these same frequencies of the impulse response $g_{v,k}$ likewise determined by L_v . Consequently, the impulse response q_k^1 of feedback filter 33 in Fig. 4 will extend over a considerably smaller time span and/or have considerably smaller amplitude values than the impulse response q_k of feedback filter 33 in Fig. 3, thus considerably reducing the risk of error propagation.

It will be evident that this advantage of reduced error propagation in data receiver 3 as shown in Fig. 4 is maintained if instead of virtual data signal c_k the actually transmitted data signal b_k is reconstructed by symbol decision circuit 32. Even then the configuration as shown in Fig. 3, in which symbol decisions \hat{b}_k with respect to actually transmitted data signal b_k are applied directly to feedback filter 33, could, according to the just described mechanism, lead to an impulse response q_k of feedback filter 33 extending over a large time span and/or having large amplitude values. Thus, serious error propagation may occur. In the configuration as shown in Fig. 4 the corresponding impulse response q_k^1 of feedback filter 33 leads, under the same circumstances, to a considerably smaller error propagation owing to the convolution of impulse response q_k and the impulse response $g_{t,k}$ corresponding with linear signal transformation L_t which is performed in linear part 13 of encoder 11 in data transmitter 1.

As explained hereinbefore, the advantages of a simplified implementation of feedback filter 33 and reduced error propagation realized by means of the configuration of Fig. 4 apply both in the case where at the input of symbol decision circuit 32 an estimate \tilde{c}_k of virtual data signal c_k is formed and in the case where an estimate \tilde{b}_k of actually transmitted data signal b_k is formed. Since these data signals c_k and b_k , respectively, are related to data signal a_k at the input of the linear part (13,21,22) of the transmission path via the linear signal transformations $L_v = L_c \bullet L_t$ and L_t , respectively, it is evident that said two advantages generally occur if at the input of symbol decision circuit 32 an estimate is formed of an m-level data signal that is related to n-level data signal a_k according to a linear signal transformation L with $L = L_v$ or $L = L_t$, which linear signal transformation L corresponds with a partial-response polynomial $g(D) = g_v(D)$ and $g(D) = g_t(D)$, respectively.

An additional advantage of the configuration of data receiver 3 shown in Fig. 4 relates to the option of adaptively implementing feedback filter 33 and possibly also feedforward filter 31. This option is illustrated in Fig. 5.

In Fig. 5, both filters 31 and 33 now comprise an adaptation circuit 31(a) and 33(a), respectively, arranged according to conventional techniques. These adaptation circuits 31(a) and 33(a) are controlled by the same error signal ϵ_k which is representative of the difference between input signal \tilde{c}_k of symbol decision circuit 32 and a data signal \hat{b} . This data signal \hat{b} is derived in a simple way from the input signal \hat{a}_k of feedback filter 33 by means of a partial-response circuit 38 in which the desired partial-response

transformation L_v is effected. By means of a difference circuit 39 the difference Δ_k between the signals \tilde{c}_k and \hat{c}_k' is formed, and in Fig. 5 this difference Δ_k is used directly as error signal ϵ_k . As is well known, in adaptive filters prescribed functions of Δ_k , such as, for example, strongly quantized versions of Δ_k , can be used as error signal ϵ_k in order to simplify their digital implementation. When using the error signal ϵ_k thus obtained it is achieved in a simple manner that, after convergence of the adaptive filters 31 and 33, the data component $\tilde{c}_k - (n \cdot w)_k$ of the input signal \tilde{c}_k of symbol decision circuit 32 is related in the desired manner to the data signal a_k at the output of precoder 12 in data transmitter 1, that is to say, via the desired linear signal transformation L_v embedded in partial-response circuit 38. The apparently more obvious implementation, in which the in and output signals \tilde{c}_k and \hat{c}_k of symbol decision circuit 32 are used directly for forming the error signal ϵ_k , true enough, also results in a linear relationship between the data component $\tilde{c}_k - (n \cdot w)_k$ of \tilde{c}_k and the data signal a_k after adaptation of filters 31 and 33, but inevitably leads to the problem that it cannot be predicted a priori which linear relationship exactly will be established, so that an undesired adjustment of equalizer 30 cannot be precluded in advance.

It is evident that the latter advantage of a predictable convergence behaviour is maintained if a desired linear signal transformation $L = L_t$ instead of a desired linear signal transformation $L = L_v$ is performed in partial-response circuit 38. As already explained hereinbefore, this linear signal transformation $L = L_t$ leads to symbol decisions \hat{b}_k of actually transmitted data signal b_k , so that in this case decoder 35 has to perform a memoryless inverse mapping $L^{-1} \bullet NL^{-1} = L_t^{-1} \bullet NL_t^{-1}$, whilst precoder 37 has to carry out the associated non-linear signal transformation $NL = NL_t$.

The predictable convergence behaviour of feedforward filter 31 and feedback filter 33 which is guaranteed by partial-response circuit 38 in Fig. 5 leads to an input signal \tilde{c}_k of symbol decision circuit 32 with a correlation structure substantially corresponding with the well-defined correlation structure of output signal \hat{c}_k' of partial-response circuit 38, which correlation structure can be characterized by a partial-response polynomial $g_v(D)$ or $g_t(D)$. This well-defined correlation structure of input signal \tilde{c}_k of symbol decision circuit 32 in Fig. 5 can now be used for realizing a further improvement of transmission quality by adding a non-adaptive post-detector 40 for forming final symbol decisions \hat{d}_{k-M} which are applied to a data signal sink 36', as shown in Fig. 5 by way of a dashed line. Such a post-detector is known from an article "Maximum-Likelihood Sequence Estimation of Digital Sequences in the Presence of Intersymbol Interference" by G.D. Forney, Jr., published in IEEE Trans. Inform. Theory, Vol. IT-18, No 3, pp. 363-378, May 1972. In this article a non-adaptive detector is described which is arranged for estimating the maximum-likelihood sequence of transmitted data symbols d_k and thereto makes optimum use of the correlation structure of its input signal \tilde{c}_k . This leads to a transmission quality which is better than when making symbol-by-symbol decisions as performed in symbol decision circuit 32. For correlation structures of the partial-response type considered, according to the article by Forney improvements of transmission quality corresponding with an improvement of 2-3 dB in the signal-to-noise ratio are often obtainable in this manner. In addition, the implementation of non-adaptive post-detector 40 can remain relatively simple as a result of the low order and the resulting short memory span of the partial-response polynomial ($g_v(D)$ or ($g_t(D)$)) which determines the correlation structure of input signal \tilde{c}_k of post-detector 40. Needless to observe that also different types of non-adaptive detectors exploiting this well-defined correlation structure for achieving an improved transmission quality can be used as post-detector 40 in Fig. 5. It will also be evident after the above explanation that in the non-adaptive configurations of the Figs. 2, 3 and 4 it may be sensible to connect such a non-adaptive post-detector to the input of symbol decision circuit 32 for forming final symbol decisions \hat{d}_{k-M} having a better quality than the decisions \hat{d}_k , but for simplicity, this option has not been shown any further in these Figures.

Claims

1. A system for transmitting an n-level data signal at a given symbol rate $1/T$, this system being constituted by a data transmitter with a data signal source, a transmission channel and a data receiver with an equalizer of the decision feedback type, which equalizer includes a feedforward filter connected between the input of the data receiver and a first input of a difference circuit, a symbol decision circuit connected to the output of the difference circuit, and a feedback filter connected between the output of the symbol decision circuit and a second input of the difference circuit, in which the linear part of the transmission path between the output of the data signal source and the first input of the difference circuit can be described by a linear signal transformation L , characterized in that the equalizer is arranged for forming at the input of the symbol decision circuit an estimate of a virtual m-level data signal related to the n-level data signal

applied to the input of the linear part of the transmission path according to a linear signal transformation L_v which substantially characterizes the linear signal transformation L and corresponds with a partial-response polynomial $g_v(D)$ with D being a delay operator representing the symbol interval T .

2. A system as claimed in Claim 1, characterized in that the data transmitter comprises a precoder
5 connected between the data signal source and the input to the linear part of the transmission path for performing a non-linear signal transformation NL_v which is unambiguously determined by the linear signal transformation L_v , and in that the feedback filter in the data receiver is connected to the output of the symbol decision circuit through a decoder and a precoder which is identical with the precoder in the data transmitter, said decoder performing a memoryless inverse signal transformation $L_v^{-1} \bullet NL_v^{-1}$ which converts
10 the m -level symbol decisions into a n -level data signal corresponding with the original n -level data signal.

3. A data receiver comprising a decision feedback equalizer which includes a feedforward filter connected between the input of the data receiver and the first input of a difference circuit, a symbol decision circuit connected to the output of the difference circuit, and a feedback filter connected between
15 the output of the symbol decision circuit and a second input of the difference circuit, said data receiver being suitable for use in a system as claimed in Claim 1 or 2 and arranged for recovering an original n -level data signal from the output signal of the feedforward filter incorporated in a linear transmission path, to which an n -level data signal having a given symbol rate $1/T$ and being derived from the original n -level data signal is applied, characterized in that the equalizer is arranged for forming at the input of the symbol decision circuit an estimate of an m -level data signal related to the derived n -level data signal according to
20 a linear signal transformation L which corresponds with a partial-response polynomial $g(D)$ with D being a delay operator representing the symbol interval T , and in that the feedback filter is connected to the output of the symbol decision circuit through a decoder and a precoder, said decoder performing a memoryless inverse signal transformation $L^{-1} \bullet NL^{-1}$ which converts the m -level symbol decisions into an n -level data signal corresponding with the original n -level data signal, NL being a non-linear signal transformation which
25 is unambiguously determined by the linear signal transformation L , and said precoder performing the non-linear signal transformation NL which converts the n -level data signal at the output of the decoder into an n -level data signal corresponding with the derived n -level data signal.

4. A data receiver comprising a decision feedback equalizer which includes a feedforward filter connected between the input of the data receiver and the first input of a difference circuit, a symbol
30 decision circuit connected to the output of the difference circuit, and a feedback filter connected between the output of the symbol decision circuit and a second input of the difference circuit, said data receiver being suitable for use in a system as claimed in Claim 1 or 2 and arranged for recovering an original n -level data signal from the output signal of the feedforward filter incorporated in a linear transmission path, to which an n -level data signal having a given symbol rate $1/T$ and being derived from the original n -level data
35 signal is applied, characterized in that the equalizer is arranged for forming at the input of the symbol decision circuit an estimate of an m -level data signal related to the derived n -level data signal according to a linear signal transformation L which corresponds with a partial-response polynomial $g(D)$ with D being a delay operator representing the symbol interval T , and in that the input signal of the symbol decision circuit is also applied to a post-detector for forming a final estimate of the original n -level data signal.

40 5. A data receiver as claimed in Claim 3, characterized in that the input signal of the symbol decision circuit is also applied to a post-detector for forming a final estimate of the original n -level data signal.

6. A data receiver as claimed in Claim 3, 4 or 5, characterized in that the feedback filter is adaptively adjustable under control of an error signal which is representative of the difference between the input signal of the symbol decision circuit and a signal derived from the input signal of the feedback filter by performing
45 the linear signal transformation L .

7. A data receiver as claimed in Claim 3, 4 or 5, characterized in that both the feedforward filter and the feedback filter are adaptively adjustable under control of an error signal which is representative of the difference between the input signal of the symbol decision circuit and a signal derived from the input signal of the feedback filter by performing the linear signal transformation L .

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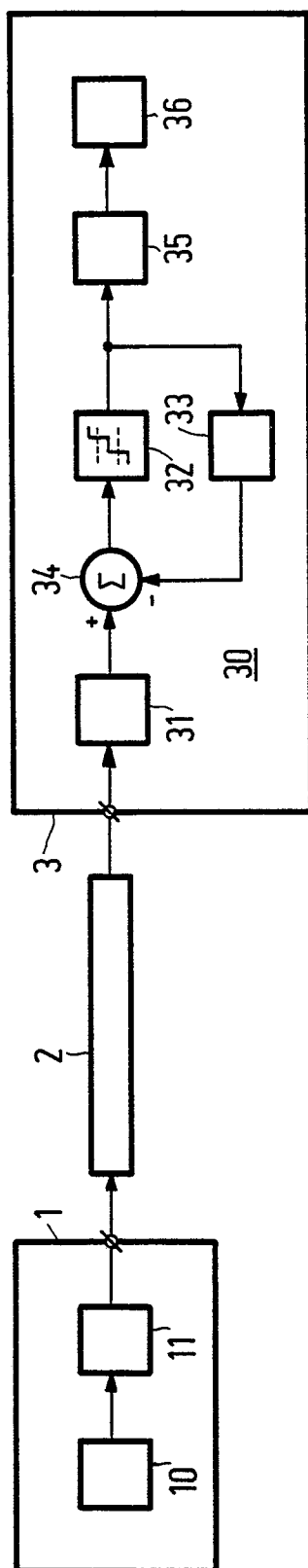


FIG.1

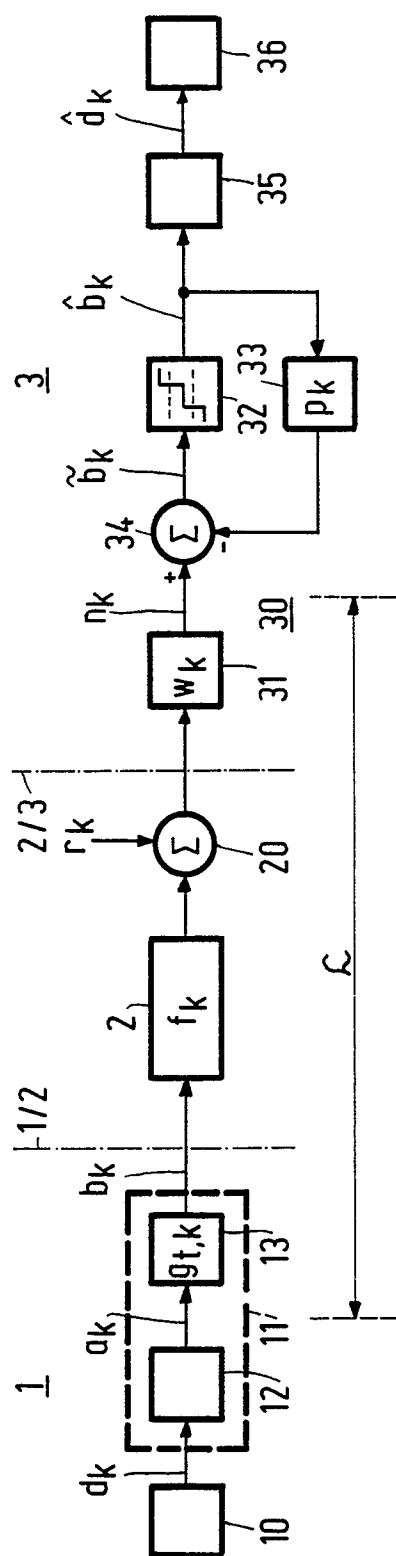


FIG. 2

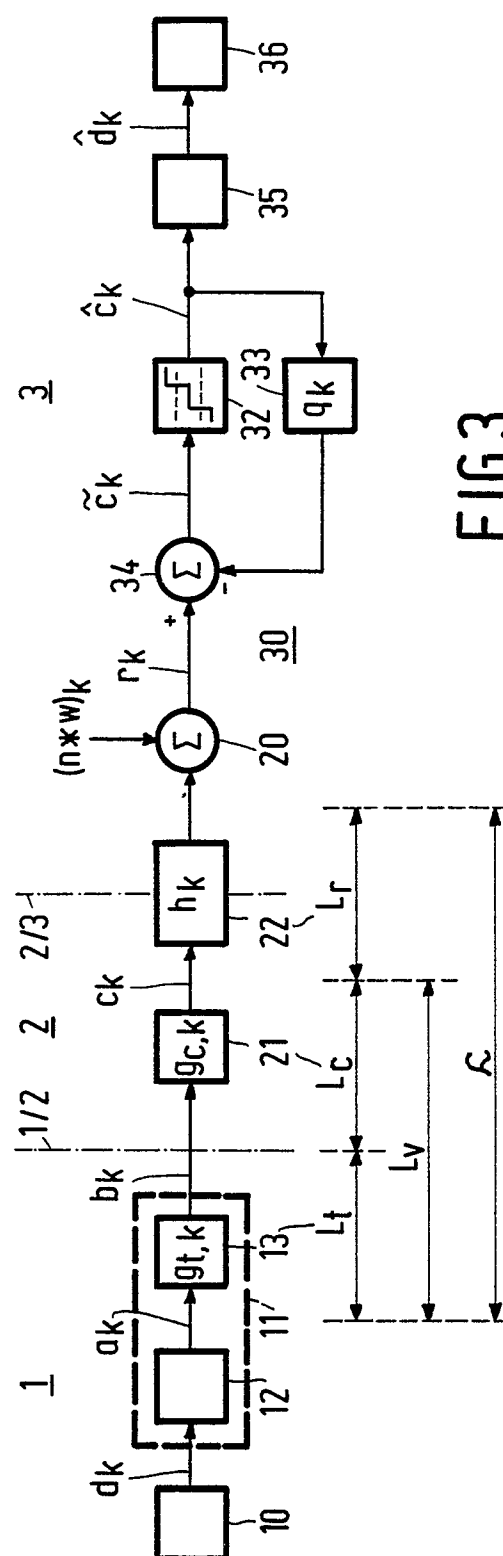


FIG. 3

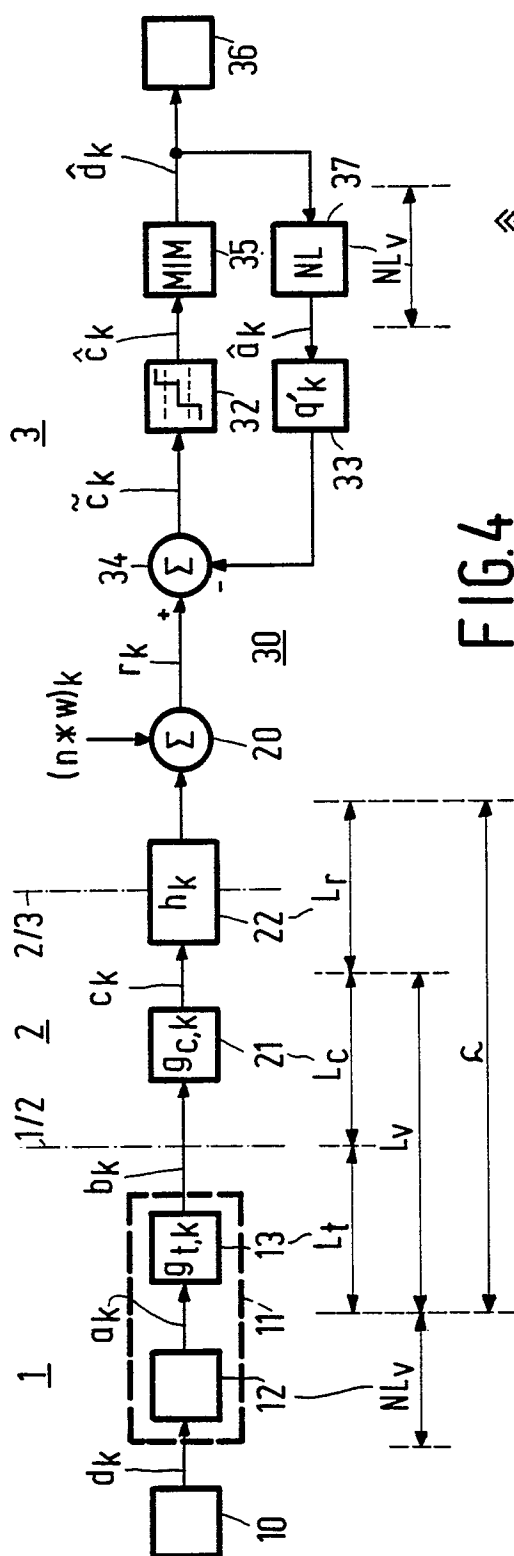


FIG. 4

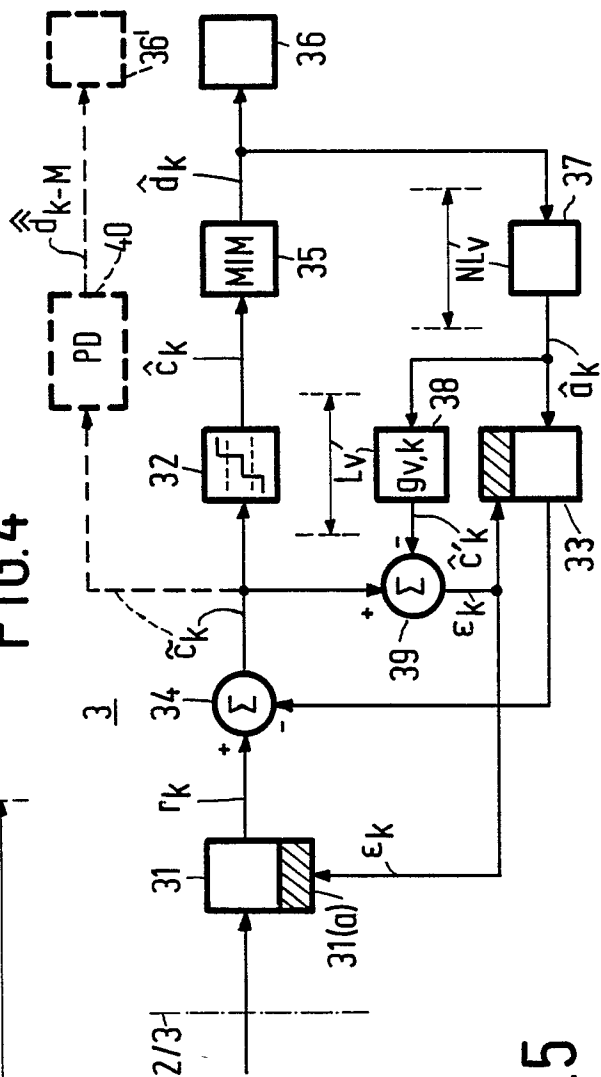


FIG. 5



DOCUMENTS CONSIDERED TO BE RELEVANT			
Category	Citation of document with indication, where appropriate, of relevant passages	Relevant to claim	CLASSIFICATION OF THE APPLICATION (Int. Cl.4)
A	NACHRICHTENTECHNIK ELEKTRONIK, vol. 33, no. 6, 1983, pages 256-261, East Berlin, DD; G. UHLRICH: "Vergleich einiger Partial-Response-Übertragungsverfahren bei PCM-Übertragung mit hohen Bitraten" * Paragraph 1 *	1-3	H 04 L 25/03
A	IEEE TRANSACTIONS ON COMMUNICATIONS, vol. COM-28, no. 8, August 1980, pages 1425-1430, IEEE, New York, US; T. ROSTE et al.: "A 140 Mbit/s digital transmission system for coaxial cable using partial response class 1 line code with quantized feedback" * Paragraph III *	1,3,4	
A	US-A-3 388 330 (KRETZMER) * Column 11, line 52 - column 12, line 2; column 12, lines 34-37 *	1,2	
A	US-A-3 648 171 (HIRSCH) * Figure 4 *	1,4-7	TECHNICAL FIELDS SEARCHED (Int. Cl.4)
A	SECOND EUROPEAN CONFERENCE ON OPTICAL FIBRE COMMUNICATION, Paris, 27th-30th September 1976, pages 333-341, Oliver de Serres, FR; R. DOGLIOTTI et al.: "Signal processing in digital fiber communications" * Figure 6 *	1,4,5	H 04 L
The present search report has been drawn up for all claims			
Place of search THE HAGUE		Date of completion of the search 19-08-1988	Examiner GEISLER J.A.R.
CATEGORY OF CITED DOCUMENTS X : particularly relevant if taken alone Y : particularly relevant if combined with another document of the same category A : technological background O : non-written disclosure P : intermediate document I : theory or principle underlying the invention E : earlier patent document, but published on, or after the filing date D : document cited in the application L : document cited for other reasons & : member of the same patent family, corresponding document			



DOCUMENTS CONSIDERED TO BE RELEVANT			
Category	Citation of document with indication, where appropriate, of relevant passages	Relevant to claim	CLASSIFICATION OF THE APPLICATION (Int. Cl. 4)
A	GLOBECOM 85, CONFERENCE RECORD OF THE IEEE GLOBAL TELECOMMUNICATIONS CONFERENCE, New Orleans, 2nd-5th December 1985, vol. 3, pages 43.3.1 - 43.3.7, IEEE, New York, US; N.-S. LIN et al.: "Partial response coding in digital subscriber loops" * Paragraph 3.3 *	1-7	
A	US-A-3 792 356 (KOBAYASHI) * Figure 2 *	1-7	
			TECHNICAL FIELDS SEARCHED (Int. Cl. 4)
The present search report has been drawn up for all claims			
Place of search THE HAGUE		Date of completion of the search 19-08-1988	Examiner GEISLER J.A.R.
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